1. Amendments to the Claims:

A listing of the entire set of pending claims (including amendments to the claims, if any) is submitted herewith per 37 CFR § 1.121. This listing of claims will replace all prior versions, and listings, of claims in the application.

Listing of Claims:

- 1. (Currently amended) In A method for correcting impairments on information, passing through an information transmission system, comprising imposed by a plurality of defective elements of the information transmission system for generating, transporting, and receiving the information, wherein some elements are defective and impose impairments on the information passing therethrough, a method for correcting said impairments, comprising:
- a) identifying the defective elements imposing impairments on the information and characterizing the each defect of each, including by performing a frequency analysis of each defective element;
- b) determining a correction characteristic corresponding to each defective element, and creating a frequency characteristic complementary to said frequency analysis, such that the a combination of said frequency analysis and said complementary frequency characteristic along with the correction characteristic, when applied to information passing through said element, corrects the impairment imposed by said element; and
- e) creating a composite, two channel I and Q finite impulse response filter, having I-I and Q-Q direct components and I-Q and Q-I cross components, by combining said correction complementary frequency characteristics, and performing an inverse discrete Fourier transform of said complementary characteristic; and g) positioning, said filter being positioned in said information transmission system for correcting said impairments imposed on the information by said defective elements.

- 2. (Currently amended) An information transmission system, as in The method of claim 1, wherein said system is limited to a data receiver whose and said plurality of elements include an IF filter, a two-channel down-converter, and I and Q data processing channels.
- 3. (Currently amended) An information transmission system, as in The method of claim 1, wherein said system is limited to a data generator whose and said plurality of elements include I and Q data channels, a two-channel up-converting modulator, and an IF filter.
 - 4. (Canceled)
- 5. (Currently amended) An information transmission system, as in The method of claim 1, wherein:
- i. step (c) further creating a composite, two channel I and Q finite impulse response filter includes arranging said direct and said cross components as terms of a set of 2x2 matrices; and
- ii. step (d) further includes arranging said single correction characteristic as terms of a set of 2x2 matrices.
- 6. (Currently amended) A generalized digital filter for filtering two-component signal information of a receiver comprising a plurality of receiver elements, the filter comprising:
- a) a dual input port, having an I input for a signal x_I and a Q input for a signal x_Q , wherein x_I and x_Q are components of a two-component input signal x;
- b) a dual output port, having an I output for a signal y_I and a Q output for a signal y_O , wherein y_I and y_O are components of a two-component output signal y_I ;
 - e) a first signal path, characterized by a first impulse response, having an input

Application Serial Number 09/846,138 Response to Office Action Dated August 11, 2008

coupled to the I input port and a first output;

- d) a second signal path, characterized by a second impulse response, having an input coupled to the Q input port and a second output;
- e) a third signal path, characterized by a third impulse response, having an input coupled to the I input port and a third output;
- f) a fourth signal path, characterized by a fourth impulse response, having an input coupled to the Q input port and a fourth output;
- g) summing means a first adder for adding said first and second outputs and for coupling the sum thereof to said I output; and
- h) summing means a second adder for adding said third and fourth outputs and for coupling the sum thereof to said Q output.

wherein said first, second, third and fourth impulse responses provide frequency compensation for impairments imposed by a plurality of defective elements among the plurality of receiver elements.

- 7. (Currently amended) A <u>The</u> generalized digital filter, as in claim 6, wherein said first, second, third, and fourth impulse responses are independent of one another.
- 8. (Currently amended) A The generalized digital filter, as in claim 7, wherein said first, second, third, and fourth impulse responses are further constrained to have finite lengths.
- 9. (Currently amended) A <u>The</u> generalized digital filter, as in claim 8, wherein said first, second, third, and fourth impulse responses are further constrained to have equal lengths.
- 10. (Currently amended) A <u>The</u> A generalized digital filter, as in claim 6, wherein said first, second, third, and fourth signal paths are realized by finite impulse-response filters.

- 11. (Currently amended) A <u>The</u> generalized digital filter, as in claim 10, wherein each of said finite impulse-response filters is independently characterized.
- 12. (Previously Presented) In applying a generalized two-channel digital filter to process an input data stream x and to produce an output data stream y, wherein both x and y are two-component signals x_I , x_Q , y_I , and y_Q which are processed in blocks of N/2 data values long, N being a power of 2, and wherein the filter is characterized by four independent impulse response vectors h_{11} , h_{12} , h_{21} , and h_{22} , each vector of length N/2, a method for efficiently computing said output data stream y, comprising the preliminary steps of:
 - a) forming the vectors

$$a = \frac{(h_{11} + h_{22}) + j(h_{21} - h_{12})}{2} \quad \text{and} \quad b = \frac{(h_{11} - h_{22}) + j(h_{21} + h_{12})}{2}$$

- b) appending N/2 zeros to each vector and performing an FFT on each vector to produce A_k and B_k , respectively; and, for each block of N/2 data values in said input data stream x, additionally comprising the iterative steps of:
- c) moving the previous block of input data values to the first half of an input vector x_N of length N and loading the current block of input data values into the second half of said input vector x_N ;
- d) treating x_N as a vector of complex numbers of the form $x_I + jx_Q$, and performing a N-point FFT to produce X_k ;
- e) computing the complex vector $Y_k = A_k X_k + B_k X_{N-k}$, $0 \le k < N/2$, and performing an inverse FFT on the result to produce the complex vector y_n ;
- f) designating the second half of y_n as the N/2 output samples of the current iteration, according to $y_{In} = \text{Real }(y_O)$, $y_n = \text{Imag }(y_n)$, where N/2 \leq n \leq N; and
 - g) returning to step (c) for the next block of N/2 data values.